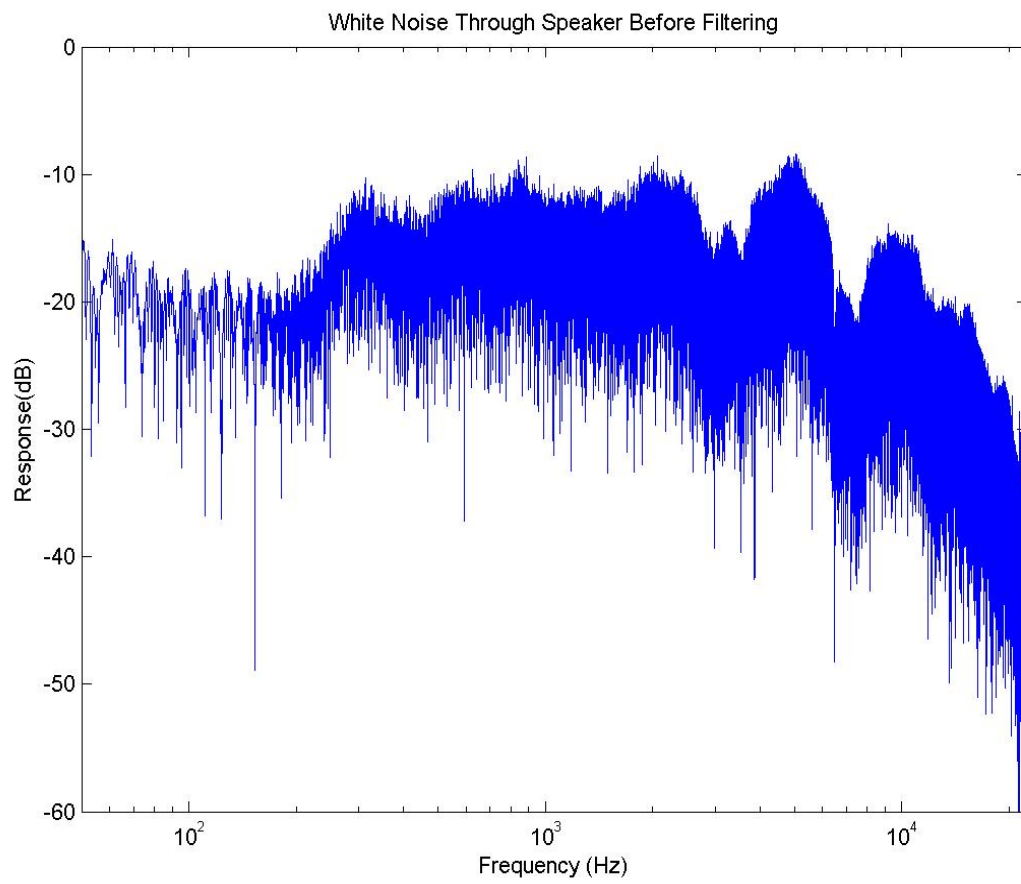Frequency Response
Elec 301 Project procedure part 1: Obtaining the Frequency Response

Procedure:

1. Frequency response:

The frequency response of the speaker describes everything from low
frequency resonance to high frequency distortion. One of the ways to find a
frequency response of a speaker is to blast white noise through the speaker
and record the result. White noise contains all frequencies and thus can be
used to describe the speaker's response to any input signal. To obtain the
frequency response of a speaker, generate white noise and record the
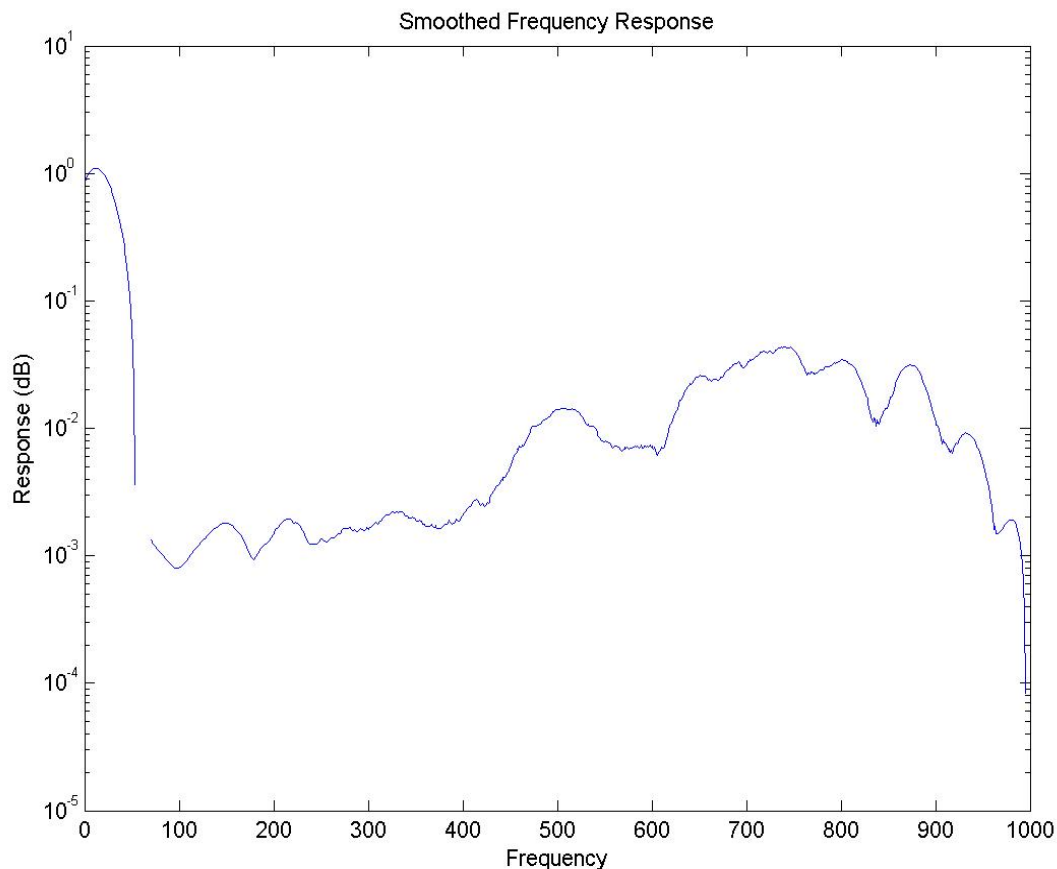response and then use Matlab's fft command and a log log plot to view the
response.

The resulting frequency response describes the speaker but is very noisy.

2. Sampling and Smoothing the Response.

Next sample the frequency response logarithmically so that low frequencies
will be weighted more importantly by the filter. This allows the logarithmic
plot of an ideal inverse filter and the inverse filter implemented by FIR
coefficients to match (rather than matching only high frequencies). Next use
MATLAB's built in Savitzky Golay [y = sgolayfilt(x,k,f)] filter to smooth
the curve so as to reduce the noise of the response. The motivation behind
this was is to prevent the fir2 algorithm from trying to create an inverse
filter with the inverse of the noise of the frequency response.

```matlab
sampfft=thisfft(round(10.^(linspace(0,log10(length(thisfft)),1000))));
%sample fft logrithmicly at 1000 points

smoothfft=sgolayfilt(sampfft, 3, 71);

smoothfft(smoothfft<0)=0; %fix smoothing errors

semilogy(smoothfft)

title('Smoothed Frequency Response');

xlabel('Frequency');

ylabel('Response (dB)');
```
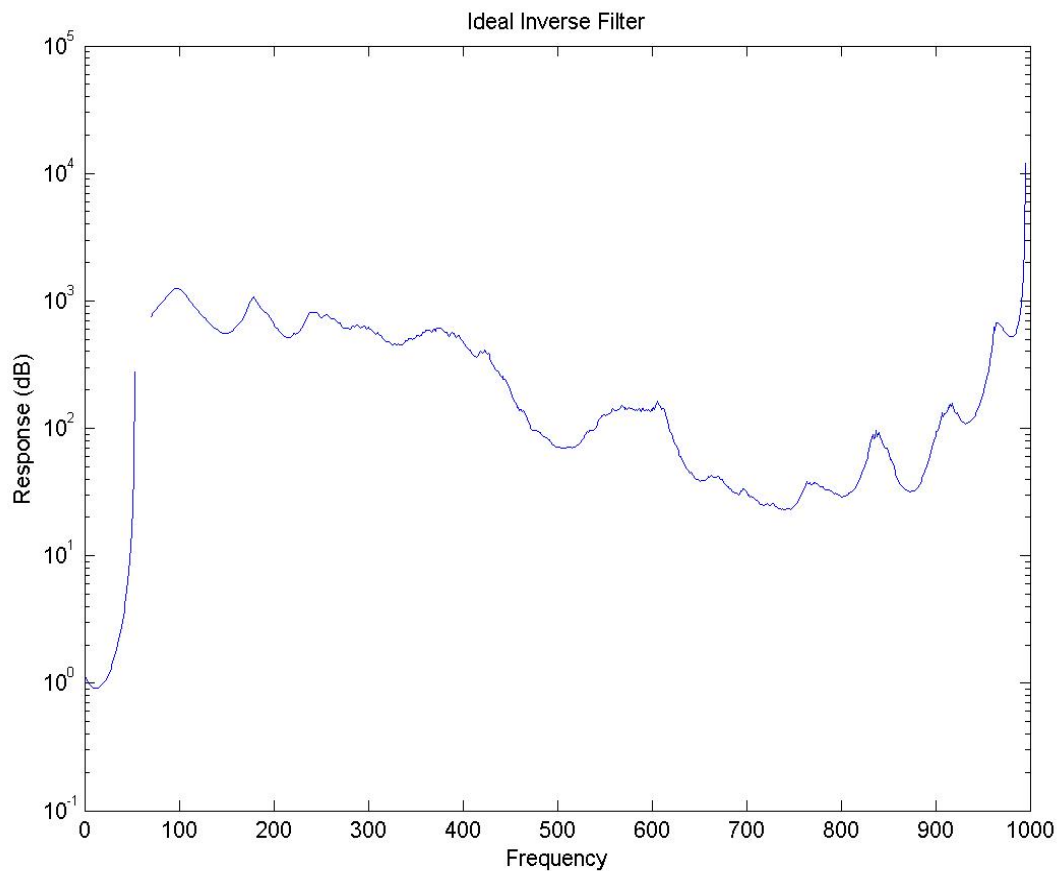
3. Low Frequency Resonance:

The resonant frequency of a loudspeaker is the frequency at which it is most susceptible to an electric response and any departure from this frequency causes the response to drop sharply. Because it is difficult to flatten such large peaks and valleys located next to one another, it's necessary to highpass filter the input to remove all frequencies at and below the resonant frequency. In the exemplar speaker, the resonant frequency zone was identified by a small bump followed by sharp drop in the frequency response at ~500 Hz.

4. Inverting the Response

After smoothing and logarithmically sampling the frequency response of a speaker one only has to invert this data to create an ideal inverse filter for the speaker. Simply flip the frequency response of the loudspeaker over the frequency axis and scale it appropriately.

targetfft=(1./smoothfft);

semilogy(targetfft)

```
title('Ideal Inverse Filter');

xlabel('Frequency');

ylabel('Response (dB)');
```

5. FIR Filter on the TI Chip.
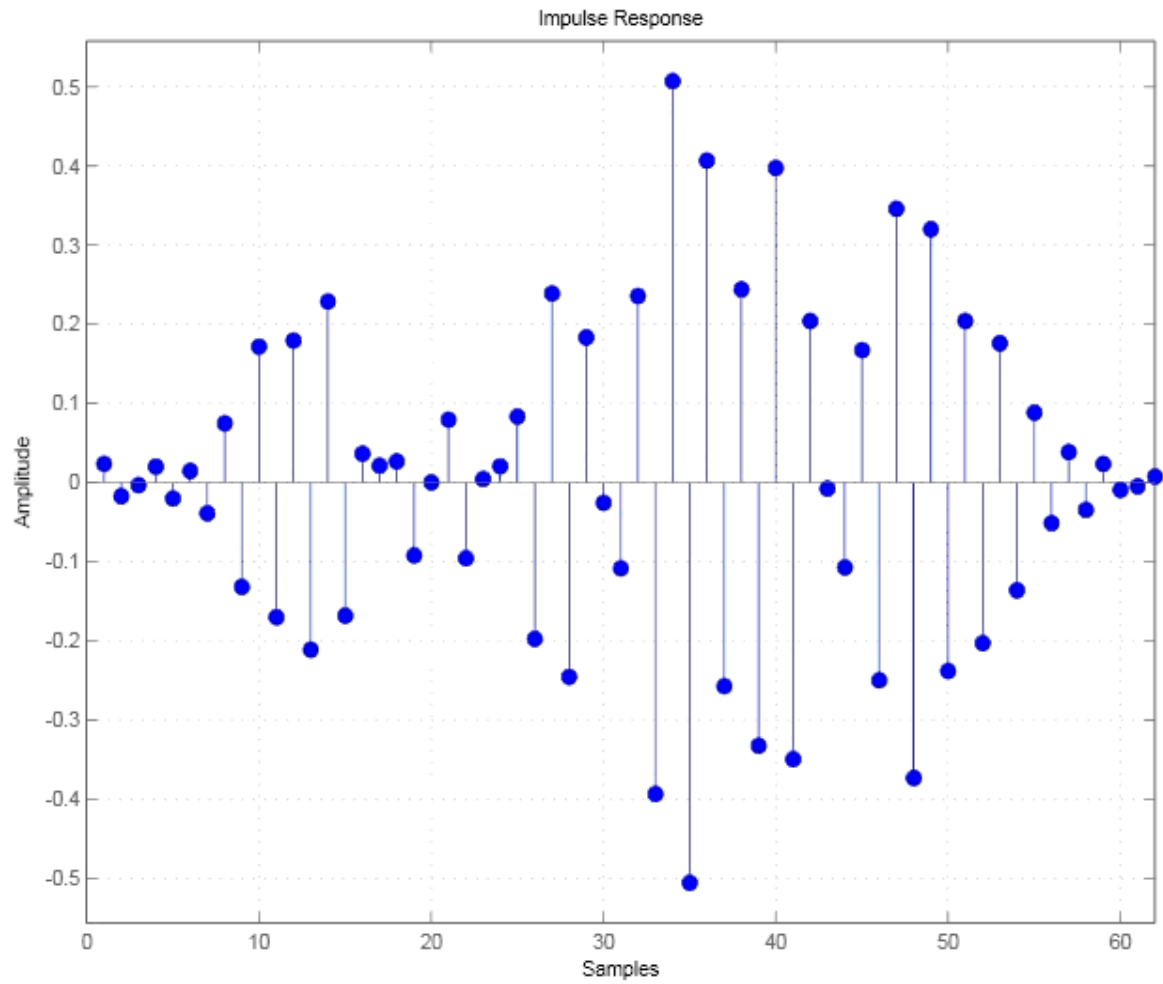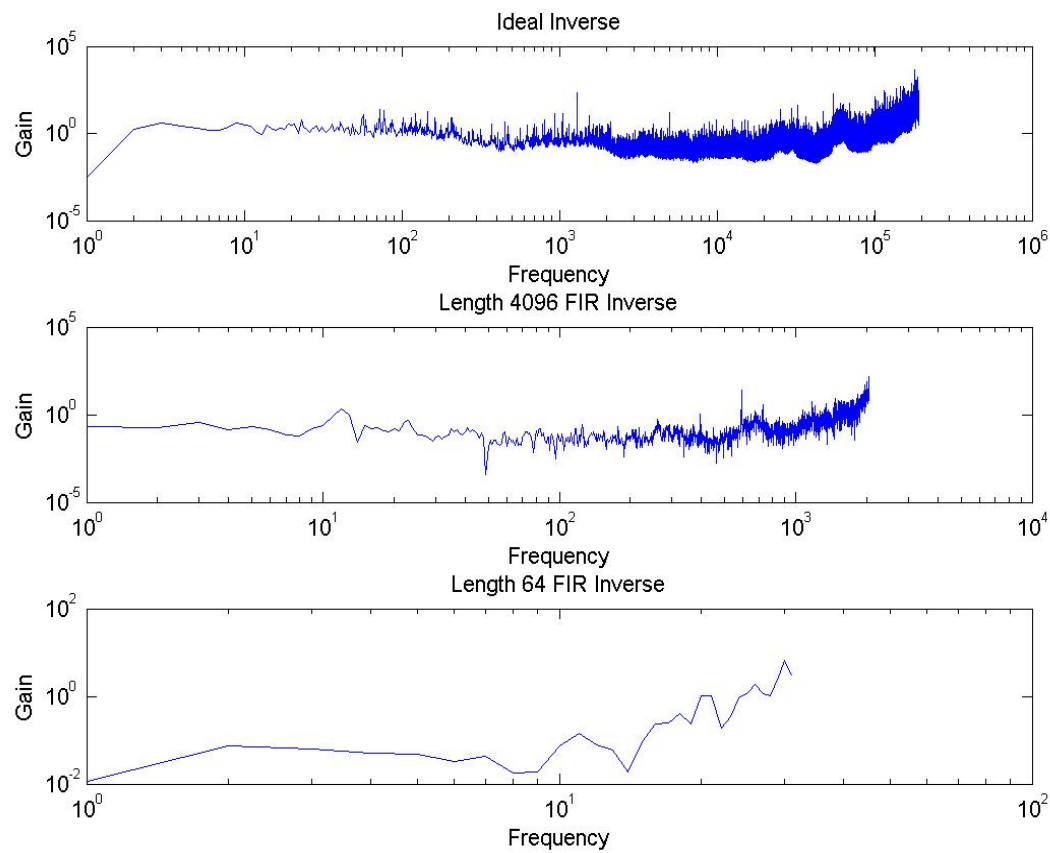
To create a FIR filter which matches a desired inverse filter, use the built in MATLAB filter design program called fir2 [b = fir2(n,f,m)]. This designs an n order FIR filter which attempts to create an FIR filter whose spectrum matches a linear interpolation between input amplitudes m at frequencies f. The coefficients (b) are obtained by applying an inverse Fourier transform to m at frequencies f and multiplying by a window. In creating an inverse filter, the default Hamming window is acceptable because it provides a balance between the dynamic range and the resolution of the signal. The response of the inverse filter at different frequencies should not be so different that a huge dynamic range is required. Finally enter these coefficients into the TI chip and produce an FIR filter.

Impulse Response

h=fir2(62, ((10.^(linspace(0,1,length(targetfft))'))-1)/9,targetfft);

Ideal Inverse

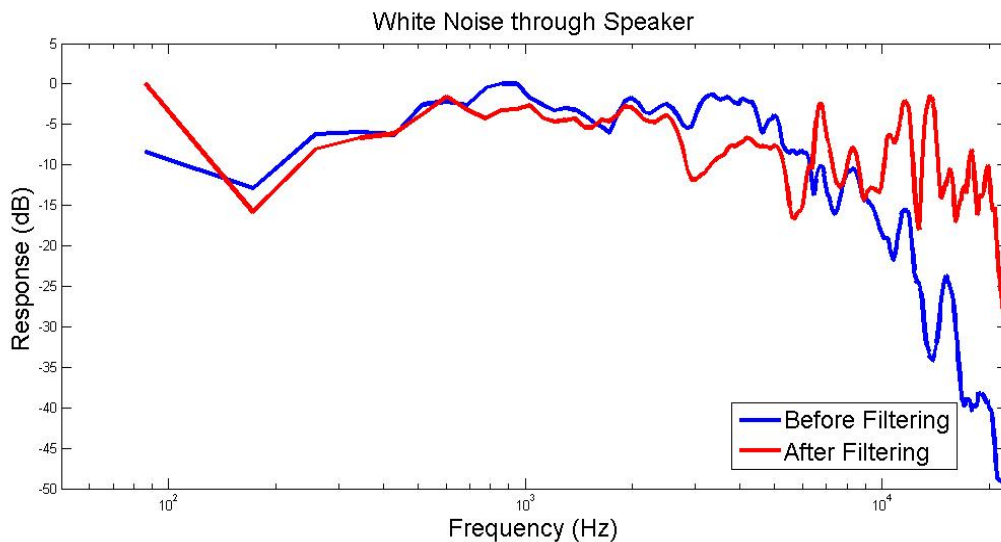**Length 4096 FIR Inverse**

Length 64 FIR Inverse

The FIR2 command with 64 coefficients does an acceptable job of matching the shape of an ideal inverse filter. However, given more memory and more coefficients one could do any even better job of matching the ideal.

6. Verification

Finally to verify that the inverse filter works, play white noise through the speaker again and record the response with and without the filter.

The filter isn't able to completely flatten the frequency response but it does smooth it significantly, especially at higher frequencies.